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Best Current Practice for Communications Services in support of Emergency Calling draft-ietf-ecrit-phonebcp-16

Abstract

The IETF and other standards organization have efforts targeted at standardizing various aspects of placing emergency calls on IP networks. This memo describes best current practice on how devices, networks and services should use such standards to make emergency calls.

Status of this Memo

This Internet-Draft is submitted in full conformance with the provisions of BCP 78 and BCP 79.

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1. Terminology

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [\[RFC2119\] \(Bradner, S., "Key words for use in RFCs to Indicate Requirement Levels," March 1997.\)](#).

This document uses terms from [\[RFC3261\] \(Rosenberg, J., Schulzrinne, H., Camarillo, G., Johnston, A., Peterson, J., Sparks, R., Handley, M., and E. Schooler, "SIP: Session Initiation Protocol," June 2002.\)](#), [\[RFC5012\] \(Schulzrinne, H. and R. Marshall, "Requirements for Emergency Context Resolution with Internet Technologies," January 2008.\)](#) and [\[I-D.ietf-ecrit-framework\] \(Rosen, B., Schulzrinne, H., Polk, J., and A. Newton, "Framework for Emergency Calling using Internet Multimedia," July 2010.\)](#).

2. Introduction

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This document describes how access networks, Session Initiation Protocol [\[RFC3261\] \(Rosenberg, J., Schulzrinne, H., Camarillo, G., Johnston, A., Peterson, J., Sparks, R., Handley, M., and E. Schooler, "SIP: Session Initiation Protocol," June 2002.\)](#) user agents, proxy servers and PSAPs support emergency calling, as outlined in [\[I-D.ietf-ecrit-framework\] \(Rosen, B., Schulzrinne, H., Polk, J., and A. Newton, "Framework for Emergency Calling using Internet Multimedia," July 2010.\)](#), which is designed to complement the present document in section headings, numbering and content. This BCP succinctly describes the requirements of end devices and applications (requirements prefaced by "ED-"), access networks (including enterprise access networks) (requirements prefaced by "AN-"), service providers (requirements prefaced by "SP-") and PSAPs to achieve globally interoperable emergency calling on the Internet.

This document also defines requirements for "Intermediate" devices which exist between end devices or applications and the access network. For example, a home router is an "Intermediate" device. Reporting location on an emergency call (see [Section 6 \(Location and its role in an emergency call\)](#)) may depend on the ability of such intermediate devices to meet the requirements prefaced by "INT-".

3. Overview of how emergency calls are placed

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An emergency call can be distinguished ([Section 5 \(Identifying an emergency call\)](#)) from any other call by a unique Service URN [\[RFC5031\] \(Schulzrinne, H., "A Uniform Resource Name \(URN\) for Emergency and](#)

[Other Well-Known Services," January 2008.](#)), which is placed in the call set-up signaling when a home or visited emergency dial string is detected. Because emergency services are local to specific geographic regions, a caller must obtain his location ([Section 6 \(Location and its role in an emergency call\)](#)) prior to making emergency calls. To get this location, either a form of measuring (e.g., GPS) ([Section 6.2.3 \(End-system measured location information\)](#)) device location in the endpoint is deployed, or the endpoint is configured ([Section 6.5 \(End system location configuration\)](#)) with its location from the access network's Location Information Server (LIS). The location is conveyed ([Section 6.7 \(Conveying location in SIP\)](#)) in the SIP signaling with the call. The call is routed ([Section 8 \(Routing the call to the PSAP\)](#)) based on location using the LoST protocol [[RFC5222](#)] ([Hardie, T., Newton, A., Schulzrinne, H., and H. Tschofenig, "LoST: A Location-to-Service Translation Protocol," August 2008.](#)), which maps a location to a set of PSAP URIs. Each URI resolves to a PSAP or an Emergency Services Routing Proxy (ESRP), which serves a group of PSAPs. The call arrives at the PSAP with the location included in the SIP INVITE request.

4. Which devices and services should support emergency calls

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ED-1 A device or application SHOULD support emergency calling if a user could reasonably expect to be able to place a call for help with the device. Some jurisdictions have regulations governing this.

SP-1 If a device or application expects to be able to place a call for help, the service provider that supports it MUST facilitate emergency calling. Some jurisdictions have regulations governing this.

ED-2 Devices that create media sessions and exchange real-time audio, video and/or text, have the capability to establish sessions to a wide variety of addresses, and communicate over private IP networks or the Internet, SHOULD support emergency calls. Some jurisdictions have regulations governing this.

5. Identifying an emergency call

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ED-3 Endpoints SHOULD recognize dial strings of emergency calls. If the service provider always knows the location of the device, then the service provider could recognize them.

SP-2 Proxy servers SHOULD recognize emergency dial strings if for some reason the endpoint does not recognize them.

ED-4/SP-3 Emergency calls MUST be marked with a Service URN in the Request-URI of the INVITE.

ED-5/SP-4 Local dial strings MUST be recognized.

ED-6/SP-5 Devices MUST be able to be configured with the home country from which the home dial string(s) can be determined.

ED-7/SP-6 Emergency dial strings SHOULD be determined from LoST [\[RFC5222\] \(Hardie, T., Newton, A., Schulzrinne, H., and H. Tschofenig, "LoST: A Location-to-Service Translation Protocol," August 2008.\)](#). Dial Strings MAY be configured directly in the device.

AN-1 LoST servers MUST return dial strings for emergency services

ED-8 Endpoints which do not recognize emergency dial strings SHOULD send dial strings as per [\[RFC4967\] \(Rosen, B., "Dial String Parameter for the Session Initiation Protocol Uniform Resource Identifier," July 2007.\)](#).

SP-7 If a proxy server recognizes dial strings on behalf of its clients it MUST recognize emergency dial strings represented by [\[RFC4967\] \(Rosen, B., "Dial String Parameter for the Session Initiation Protocol Uniform Resource Identifier," July 2007.\)](#) and SHOULD recognize emergency dial strings represented by a tel URI [\[RFC3966\] \(Schulzrinne, H., "The tel URI for Telephone Numbers," December 2004.\)](#).

ED-9 Endpoints SHOULD be able to have home dial strings provisioned.

SP-8 Service providers MAY provision home dial strings in devices.

ED-10 Devices SHOULD NOT have one button emergency calling initiation.

ED-11/SP-9 All sub-services for the 'sos' service specified in [\[RFC5031\] \(Schulzrinne, H., "A Uniform Resource Name \(URN\) for Emergency and Other Well-Known Services," January 2008.\)](#) MUST be recognized.

6. Location and its role in an emergency call

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Handling location for emergency calling usually involves several steps to process and multiple elements are involved. In Internet emergency calling, where the endpoint is located is "determined" using a variety of measurement or wiretracing methods. Endpoints may be "configured" with their own location by the access network. In some circumstances, a proxy server may insert location into the signaling on behalf of the endpoint. The location is "mapped" to the URI to send the call to, and the location is "conveyed" to the PSAP (and other elements) in the signaling. Likewise, we employ Location Configuration Protocols, the Location-to-Service Mapping Protocol, and Location Conveyance Protocols for these functions. The Location-to-Service Translation protocol [\[RFC5222\] \(Hardie, T., Newton, A., Schulzrinne, H., and H. Tschofenig, "LoST: A Location-to-Service Translation Protocol," August 2008.\)](#) is the Location Mapping Protocol defined by the IETF.

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6.1. Types of location information

There are several forms of location. In IETF location configuration and location conveyance protocols, civic and geospatial (geo) forms are both supported. The civic forms include both postal and jurisdictional fields. A cell tower/sector can be represented as a point (geo or civic) or polygon. Other forms of location representation must be mapped into either a geo or civic for use in emergency calls.

ED-12/INT-1/SP-10 Endpoints, Intermediate Devices and Service Providers MUST be prepared to handle location represented in either civic or geo form.

ED-13/INT-2/SP-11/AN-2 Elements MUST NOT convert (civic to geo or geo to civic) from the form of location the determination mechanism supplied.

6.2. Location Determination

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ED-14/INT-3/AN-3 Any location determination mechanism MAY be used, provided the accuracy of the location meets local requirements.

6.2.1. User-entered location information

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ED-15/INT-4/AN-4 Devices, intermediate Devices and/or access networks SHOULD support a manual method to override the location the access network determines. Where a civic form of location is provided, all fields in the PIDF-LO [\[RFC4119\] \(Peterson, J., "A Presence-based GEOPRIV Location Object Format," December 2005.\)](#) and [\[RFC5139\] \(Thomson, M. and J. Winterbottom, "Revised Civic Location Format for Presence Information Data Format Location Object \(PIDF-LO\)," February 2008.\)](#) MUST be able to be specified.

6.2.2. Access network "wire database" location information

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AN-5 Access networks supporting copper, fiber or other hard wired IP packet service SHOULD support location configuration. If the network does not support location configuration, it MUST require every device that connects to the network to support end system measured location.

AN-6/INT-5 Access networks and intermediate devices providing wire database location information SHOULD provide interior location data (building, floor, room, cubicle) where possible. It is RECOMMENDED that interior location be provided when spaces exceed approximately 650

square meters. See [\[I-D.ietf-ecrit-framework\] \(Rosen, B., Schulzrinne, H., Polk, J., and A. Newton, "Framework for Emergency Calling using Internet Multimedia," July 2010.\)](#) Section 6.2.2 for a discussion of how this value was determined.

AN-7/INT-6 Access networks and intermediate devices (including enterprise networks) which support intermediate range wireless connections (typically 100m or less of range) and which do not support a more accurate location determination mechanism such as triangulation, MUST support location configuration where the location of the access point is reflected as the location of the clients of that access point.

AN-8/INT-7 Where the access network provides location configuration, intermediate devices MUST either be transparent to it, or provide an interconnected client for the supported configuration mechanism and a server for a configuration protocol supported by end devices downstream of the intermediate device such that the location provided by the access network is available to clients as if the intermediate device was not in the path.

6.2.3. End-system measured location information

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ED-16/INT-8 Devices MAY support end-system measured location. See [\[I-D.ietf-ecrit-framework\] \(Rosen, B., Schulzrinne, H., Polk, J., and A. Newton, "Framework for Emergency Calling using Internet Multimedia," July 2010.\)](#) Section 6 for a discussion of accuracy of location.

ED-17/INT-9/AN-9 Devices that support endpoint measuring of location MUST have at least a coarse location capability (typically <1km accuracy when not location hiding) for routing of calls. The location mechanism MAY be a service provided by the access network.

6.2.4. Network-measured location information

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AN-10 Access networks MAY provide network-measured location determination. Wireless access networks that do not supply network measured location MUST require that all devices connected to the network have end-system measured location. Uncertainty and confidence may be specified by local regulation. Where not specified, uncertainty of less than 100 m with 95% confidence is recommended for dispatch location.

AN-11 Access networks that provide network measured location MUST have at least a coarse location (typically <1km when not location hiding) capability at all times for routing of calls.

AN-12 Access networks with range of <10 meters (e.g. personal area networks such as Bluetooth MUST provide a location to mobile devices

connected to them. The location provided SHOULD be that of the access point location unless a more accurate mechanism is provided.

6.3. Who adds location, endpoint or proxy

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ED-18/INT-10 Endpoints SHOULD attempt to configure their own location using the LCPs listed in ED-21.

SP-12 Proxies MAY provide location on behalf of devices if:

- *The proxy has a relationship with all access networks the device could connect to, and the relationship allows it to obtain location.

- *The proxy has an identifier, such as an IP address, that can be used by the access network to determine the location of the endpoint, even in the presence of NAT and VPN tunnels that may obscure the identifier between the access network and the service provider.

ED-19/INT-11/SP-13 Where proxies provide location on behalf of endpoints, the service provider MUST ensure that either the end device is provided with the local dial strings for its current location (where the end device recognizes dial strings), or the service provider proxy MUST detect the appropriate local dial strings at the time of the call.

6.4. Location and references to location

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ED-20/INT-12 Devices SHOULD be able to accept and forward location by value or by reference. An end device that receives location by reference (and does not also get the corresponding value) MUST be able to perform a dereference operation to obtain a value.

6.5. End system location configuration

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ED-21/INT-13 Devices MUST support both the DHCP location options [\[RFC4776\] \(Schulzrinne, H., "Dynamic Host Configuration Protocol \(DHCPv4 and DHCPv6\) Option for Civic Addresses Configuration Information," November 2006.\)](#), [\[RFC3825\] \(Polk, J., Schnizlein, J., and M. Linsner, "Dynamic Host Configuration Protocol Option for Coordinate-based Location Configuration Information," July 2004.\)](#) and HELD [\[RFC5985\] \(Barnes, M., "HTTP-Enabled Location Delivery \(HELD\)," September 2010.\)](#). When devices deploy a specific access network

interface for which location configuration mechanisms such as LLDP-MED [[LLDP-MED](#)] (TIA, "ANSI/TIA-1057 Link Layer Discovery Protocol - Media Endpoint Discovery," .) or 802.11v are specified, the device SHOULD support the additional respective access network specific location configuration mechanism.

AN-13/INT-14 The access network MUST support either DHCP location options or HELD. The access network SHOULD support other location configuration technologies that are specific to the type of access network. If the access network supports more than one location configuration protocol, all such protocols MUST return the same location, within the constraints of the protocols deployed.

AN-14/INT-15 Where a router is employed between a LAN and WAN in a small (less than approximately 650 square meters) area, the router MUST be transparent to the location provided by the WAN to the LAN. This may mean the router must obtain location as a client from the WAN, and supply an LCP server to the LAN with the location it obtains. Where the area is larger, the LAN MUST have a location configuration mechanism satisfying the requirements of this document.

ED-22/INT-16 Endpoints SHOULD try all LCPs supported by the device in any order or in parallel. The first one that succeeds in supplying location MUST be used.

AN-15/INT-17 Access networks that support more than one LCP MUST reply with the same location information (within the limits of the data format for the specific LCP) for all LCPs it supports.

ED-23/INT-18/SP-14 When HELD is the LCP, the request MUST specify a value of "emergencyRouting" for the "responseTime" parameter and use the resulting location for routing. If a value for dispatch location will be sent, another request with the "responseTime" parameter set to "emergencyDispatch" must be completed, with the result sent for dispatch purposes.

ED-24 Where the operating system supporting application programs which need location for emergency calls does not allow access to Layer 2 and Layer 3 functions necessary for a client application to use DHCP location options and/or other location technologies that are specific to the type of access network, the operating system MUST provide a published API conforming to ED-12 through ED-23 and ED-25 through ED-32. It is RECOMMENDED that all operating systems provide such an API.

6.6. When location should be configured

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ED-25/INT-19 Endpoints SHOULD obtain location immediately after obtaining local network configuration information.

ED-26/INT-20 If the device is configured to use DHCP for bootstrapping, it MUST include both options for location acquisition (civic and geodetic), the option for LIS discovery, and the option for LoST

discovery as defined in [\[RFC4776\]](#) (Schulzrinne, H., "Dynamic Host Configuration Protocol (DHCPv4 and DHCPv6) Option for Civic Addresses Configuration Information," November 2006.), [\[RFC3825\]](#) (Polk, J., Schnizlein, J., and M. Linsner, "Dynamic Host Configuration Protocol Option for Coordinate-based Location Configuration Information," July 2004.), [\[RFC5986\]](#) (Thomson, M. and J. Winterbottom, "Discovering the Local Location Information Server (LIS)," September 2010.) and [\[RFC5223\]](#) (Schulzrinne, H., Polk, J., and H. Tschofenig, "Discovering Location-to-Service Translation (LoST) Servers Using the Dynamic Host Configuration Protocol (DHCP)," August 2008.).

ED-27/INT-21 If the device sends a DHCPINFORM message, it MUST include both options for location acquisition (civic and geodetic), the option for LIS discovery, and the option for LoST discovery as defined in [\[RFC4776\]](#) (Schulzrinne, H., "Dynamic Host Configuration Protocol (DHCPv4 and DHCPv6) Option for Civic Addresses Configuration Information," November 2006.), [\[RFC3825\]](#) (Polk, J., Schnizlein, J., and M. Linsner, "Dynamic Host Configuration Protocol Option for Coordinate-based Location Configuration Information," July 2004.), [\[RFC5986\]](#) (Thomson, M. and J. Winterbottom, "Discovering the Local Location Information Server (LIS)," September 2010.) and [\[RFC5223\]](#) (Schulzrinne, H., Polk, J., and H. Tschofenig, "Discovering Location-to-Service Translation (LoST) Servers Using the Dynamic Host Configuration Protocol (DHCP)," August 2008.).

ED-28/INT-22 To minimize the effects of VPNs that do not allow packets to be sent via the native hardware interface rather than via the VPN tunnel, location configuration SHOULD be attempted before such tunnels are established.

ED-29/INT-23 Software which uses LCPs SHOULD locate and use the actual hardware network interface rather than a VPN tunnel interface to direct LCP requests to the LIS in the actual access network.

AN-16 Network administrators MUST take care in assigning IP addresses such that VPN address assignments can be distinguished from local devices (by subnet choice, for example), and LISs SHOULD NOT attempt to provide location to addresses that arrive via VPN connections unless it can accurately determine the location for such addresses.

AN-17 Placement of NAT devices where an LCP uses IP address for an identifier SHOULD consider the effect of the NAT on the LCP. The address used to query the LIS MUST be able to correctly identify the record in the LIS representing the location of the querying device

ED-30/INT-24 For devices which are not expected to change location, refreshing location on the order of once per day is RECOMMENDED.

ED-31/INT-25 For devices which roam, refresh of location information SHOULD be more frequent, with the frequency related to the mobility of the device and the ability of the access network to support the refresh operation. If the device detects a state change that might indicate having moved, for example when it changes access points, the device SHOULD refresh its location.

ED-32/INT-26/AN-18 It is RECOMMENDED that location determination not take longer than 250 ms to obtain routing location and systems SHOULD

be designed such that the typical response is under 100 ms. However, as much as 3 seconds to obtain routing location MAY be tolerated if location accuracy can be substantially improved over what can be obtained in 250 ms.

6.7. Conveying location in SIP

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ED-33/SP-15 Location sent between SIP elements MUST be conveyed using [\[I-D.ietf-sip-location-conveyance\]](#) (Polk, J. and B. Rosen, "Location Conveyance for the Session Initiation Protocol," March 2009.).

6.8. Location updates

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ED-34/AN-19 Where the absolute location or the accuracy of location of the endpoint may change between the time the call is received at the PSAP and the time dispatch is completed, location update mechanisms MUST be implemented and used.

ED-35/AN-20 Mobile devices MUST be provided with a mechanism to get repeated location updates to track the motion of the device during the complete processing of the call.

ED-36/AN-21 The LIS SHOULD provide a location reference which permits a subscription with appropriate filtering.

ED-37/AN-22 For calls sent with location-by-reference, with a SIP or SIPS scheme, the server resolving the reference MUST support a SUBSCRIBE [\[RFC3265\]](#) (Roach, A., "Session Initiation Protocol (SIP)-Specific Event Notification," June 2002.) to the presence event [\[RFC3856\]](#) (Rosenberg, J., "A Presence Event Package for the Session Initiation Protocol (SIP)," August 2004.). For other location-by-reference schemes that do not support subscription, the PSAP will have to repeatedly dereference the URI to determine if the device moved.

ED-38 If location was sent by value, and the endpoint gets updated location, it MUST send the updated location to the PSAP via a SIP re-INVITE or UPDATE request. Such updates SHOULD be limited to no more than one update every 10 seconds, a value selected to keep the load on a large PSAP manageable, and yet provide sufficient indication to the PSAP of motion.

6.9. Multiple locations

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ED-39/SP-16 If the LIS has more than one location for an endpoint it MUST conform to the rules in Section 3 of [\[RFC5491\]](#) (Winterbottom, J.,

[Thomson, M., and H. Tschofenig, "GEOPRIV Presence Information Data Format Location Object \(PIDF-LO\) Usage Clarification, Considerations, and Recommendations," March 2009.\)](#)

ED-40 If a UA has more than one location available to it, it MUST choose one location to route the call towards the PSAP. If multiple locations are in a single PIDF, the procedures in [\[RFC5491\]](#) (Winterbottom, J., Thomson, M., and H. Tschofenig, "GEOPRIV Presence Information Data Format Location Object (PIDF-LO) Usage Clarification, Considerations, and Recommendations," March 2009.) MUST be followed. If the UA has multiple PIDFs, and has no reasonable basis to choose from among them, a random choice is acceptable.

SP-17 If a proxy inserts location on behalf of an endpoint, and it has multiple locations available for the endpoint it MUST choose one location to use to route the call towards the PSAP. If multiple locations are in a single PIDF, the procedures in [\[RFC5491\]](#) (Winterbottom, J., Thomson, M., and H. Tschofenig, "GEOPRIV Presence Information Data Format Location Object (PIDF-LO) Usage Clarification, Considerations, and Recommendations," March 2009.) MUST be followed. If the proxy has multiple PIDFs, and has no reasonable basis to choose from among them, a random choice is acceptable.

SP-18 If a proxy is attempting to insert location but the UA conveyed a location to it, the proxy MUST use the UA's location for routing in the initial INVITE and MUST convey that location towards the PSAP. It MAY also include what it believes the location to be in a separate Geolocation header.

SP-19 All location objects received by a proxy MUST be delivered to the PSAP.

ED-41/SP-20 Location objects MUST be created with information about the method by which the location was determined, such as GPS, manually entered, or based on access network topology included in a PIDF-LO "method" element. In addition, the source of the location information MUST be included in a PIDF-LO "provided-by" element.

ED-42/SP-21 A location with a method of "derived" MUST NOT be used unless no other location is available.

6.10. Location validation

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AN-23 A LIS should perform location validation of civic locations via LoST before entering a location in its database.

ED-44 Endpoints SHOULD validate civic locations when they receive them from their LCP. Validation SHOULD be performed in conjunction with the LoST route query to minimize load on the LoST server.

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6.11. Default location

AN-24 When the access network cannot determine the actual location of the caller, it MUST supply a default location. The default SHOULD be chosen to be as close to the probable location of the device as the network can determine. See [\[I-D.ietf-ecrit-framework\] \(Rosen, B., Schulzrinne, H., Polk, J., and A. Newton, "Framework for Emergency Calling using Internet Multimedia," July 2010.\)](#)

SP-22 Proxies handling emergency calls MUST insert a default location in the INVITE if the call does not contain a location and the proxy does not have a method for obtaining a better location.

AN-25/SP-23 Default locations MUST be marked with method=Default and the proxy MUST be identified in provided-by element of the PIDF-LO.

6.12. Other location considerations

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ED-45 If the LCP does not return location in the form of a PIDF-LO [\[RFC4119\] \(Peterson, J., "A Presence-based GEOPRIV Location Object Format," December 2005.\)](#), the endpoint MUST map the location

information it receives from the configuration protocol to a PIDF-LO.

ED-46/AN-26 To prevent against spoofing of the DHCP server, elements implementing DHCP for location configuration SHOULD use [\[RFC3118\] \(Droms, R. and W. Arbaugh, "Authentication for DHCP Messages," June 2001.\)](#) although the difficulty in providing appropriate credentials is significant.

ED-47 If S/MIME is used, the INVITE message MUST provide enough information unencrypted for intermediate proxies to route the call based on the location information included. This would include the Geolocation header, and any bodies containing location information. Use of S/MIME with emergency calls is NOT RECOMMENDED.

ED-48/SP-24 Either TLS or IPSEC [\[RFC3986\] \(Berners-Lee, T., Fielding, R., and L. Masinter, "Uniform Resource Identifier \(URI\): Generic Syntax," January 2005.\)](#) MUST be used to protect location (but see [Section 9.1 \(Use of TLS\)](#)).

7. LIS and LoST Discovery

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ED-49 Endpoints MUST support one or more mechanisms that allow them to determine their public IP address, for example, STUN [\[RFC5389\] \(Rosenberg, J., Mahy, R., Matthews, P., and D. Wing, "Session Traversal Utilities for NAT \(STUN\)," October 2008.\)](#).

ED-50 Endpoints MUST support LIS discovery as described in [\[RFC5986\] \(Thomson, M. and J. Winterbottom, "Discovering the Local Location Information Server \(LIS\)," September 2010.\)](#), and the LoST discovery as

described in [\[RFC5223\] \(Schulzrinne, H., Polk, J., and H. Tschofenig, "Discovering Location-to-Service Translation \(LoST\) Servers Using the Dynamic Host Configuration Protocol \(DHCP\)," August 2008.\)](#).

ED-51 The device MUST have a configurable default LoST server parameter. If the device is provided by or managed by a service provider, it is expected that the service provider will configure this option.

ED-52 DHCP LoST discovery MUST be used, if available, in preference to configured LoST servers. That is, the endpoint MUST send queries to this LoST server first, using other LoST servers only if these queries fail.

AN-27 Access networks which support DHCP MUST implement the LIS and LoST discovery options in their DHCP servers and return suitable server addresses as appropriate.

8. Routing the call to the PSAP

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ED-53 Endpoints who obtain their own location SHOULD perform LoST mapping to the PSAP URI.

ED-54 Mapping SHOULD be performed at boot time and whenever location changes beyond the service boundary obtained from a prior LoST mapping operation or the time-to-live value of that response has expired. The value MUST be cached for possible later use.

ED-55 The endpoint MUST attempt to update its location at the time of an emergency call. If it cannot obtain a new location quickly (see [Section 6 \(Location and its role in an emergency call\)](#)), it MUST use the cached value.

ED-56 The endpoint SHOULD attempt to update the LoST mapping at the time of an emergency call. If it cannot obtain a new mapping quickly, it MUST use the cached value. If the device cannot update the LoST mapping and does not have a cached value, it MUST signal an emergency call without a Route header containing a PSAP URI.

SP-25 Networks MUST be designed so that at least one proxy in the outbound path will recognize emergency calls with a Request URI of the service URN in the "sos" tree. An endpoint places a service URN in the Request URI to indicate that the endpoint understood the call was an emergency call. A proxy that processes such a call looks for the presence of a SIP Route header field with a URI of a PSAP. Absence of such a Route header indicates the UAC was unable to invoke LoST and the proxy MUST perform the LoST mapping and insert a Route header field with the URI obtained.

SP-26 To deal with old user agents that predate this specification and with UAs that do not have access to their own location data, a proxy that recognizes a call as an emergency call that is not marked as such (see [Section 5 \(Identifying an emergency call\)](#)) MUST also perform this mapping, with the best location it has available for the endpoint. The

resulting PSAP URI would be placed in a Route header with the service URN in the Request URI.

SP-27 Proxy servers performing mapping SHOULD use location obtained from the access network for the mapping. If no location is available, a default location (see [Section 6.11 \(Default location\)](#)) MUST be supplied.

SP-28 A proxy server which attempts mapping and fails to get a mapping MUST provide a default mapping. A suitable default mapping would be the mapping obtained previously for the default location appropriate for the caller.

ED-57/SP-29 [\[RFC3261\] \(Rosenberg, J., Schulzrinne, H., Camarillo, G., Johnston, A., Peterson, J., Sparks, R., Handley, M., and E. Schooler, "SIP: Session Initiation Protocol," June 2002.\)](#) and [\[RFC3263\] \(Rosenberg, J. and H. Schulzrinne, "Session Initiation Protocol \(SIP\): Locating SIP Servers," June 2002.\)](#) procedures MUST be used to route an emergency call towards the PSAP's URI.

9. Signaling of emergency calls

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9.1. Use of TLS

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ED-58/SP-30 Either TLS or IPsec MUST be used when attempting to signal an emergency call.

ED-59/SP-31 If TLS session establishment is not available, or fails, the call MUST be retried without TLS.

ED-60/SP-32 [\[RFC5626\] \(Jennings, C., Mahy, R., and F. Audet, "Managing Client-Initiated Connections in the Session Initiation Protocol \(SIP\)," October 2009.\)](#) is RECOMMENDED to maintain persistent TLS connections between elements when one of the element is an endpoint. Persistent TLS connection between proxies is RECOMMENDED using any suitable mechanism.

ED-61/AN-28 TLS MUST be used when attempting to retrieve location (configuration or dereferencing) with HELD. The use of [\[RFC5077\] \(Salowey, J., Zhou, H., Eronen, P., and H. Tschofenig, "Transport Layer Security \(TLS\) Session Resumption without Server-Side State," January 2008.\)](#) is RECOMMENDED to minimize the time to establish TLS sessions without keeping server-side state.

ED-62/AN-29 When TLS session establishment fails, the location retrieval MUST be retried without TLS.

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9.2. SIP signaling requirements for User Agents

ED-63 The initial SIP signaling method is an INVITE request:

1. The Request URI SHOULD be the service URN in the "sos" tree, If the device cannot interpret local dial strings, the Request-URI SHOULD be a dial string URI [\[RFC4967\] \(Rosen, B., "Dial String Parameter for the Session Initiation Protocol Uniform Resource Identifier," July 2007.\)](#) with the dialed digits.
2. The To header field SHOULD be a service URN in the "sos" tree. If the device cannot interpret local dial strings, the To: SHOULD be a dial string URI with the dialed digits.
3. The From header field SHOULD contain the AoR of the caller.
4. A Route header field SHOULD be present with a PSAP URI obtained from LoST (see [Section 8 \(Routing the call to the PSAP\)](#)). If the device does not interpret dial plans, or was unable to obtain a route from a LoST server, no such Route header field will be present.
5. A Contact header field MUST be globally routable, for example a GRUU [\[RFC5627\] \(Rosenberg, J., "Obtaining and Using Globally Routable User Agent URIs \(GRUUs\) in the Session Initiation Protocol \(SIP\)," October 2009.\)](#), and be valid for several minutes following the termination of the call, provided that the UAC remains registered with the same registrar, to permit an immediate call-back to the specific device which placed the emergency call. It is acceptable if the UAC inserts a locally routable URI and a subsequent B2BUA maps that to a globally routable URI.
6. Other header fields MAY be included as per normal SIP behavior.
7. A Supported header field MUST be included with the 'geolocation' option tag [\[I-D.ietf-sip-location-conveyance\] \(Polk, J. and B. Rosen, "Location Conveyance for the Session Initiation Protocol," March 2009.\)](#), unless the device does not understand the concept of SIP location.
8. If a device understands the SIP location conveyance [\[I-D.ietf-sip-location-conveyance\] \(Polk, J. and B. Rosen, "Location Conveyance for the Session Initiation Protocol," March 2009.\)](#) extension and has its location available, it MUST include location either by-value, by-reference or both.
9. If a device understands the SIP Location Conveyance extension and has its location unavailable or unknown to that device, it MUST include a Supported header field with a "geolocation"

option tag, and MUST NOT include a Geolocation header field, and not include a PIDF-LO message body.

10. A SDP offer SHOULD be included in the INVITE. If voice is supported the offer MUST include the G.711 codec, see [Section 14 \(Media\)](#). As PSAPs may support a wide range of media types and codecs, sending an offerless INVITE may result in a lengthy return offer, but is permitted. Cautions in [\[RFC3261\] \(Rosenberg, J., Schulzrinne, H., Camarillo, G., Johnston, A., Peterson, J., Sparks, R., Handley, M., and E. Schooler, "SIP: Session Initiation Protocol," June 2002.\)](#) on offerless INVITES should be considered before such use.
11. If the device includes location-by-value, the UA MUST support multipart message bodies, since SDP will likely be also in the INVITE.
12. A UAC SHOULD include a "inserted-by" header parameter with its own hostname on all Geolocation header fields. This informs downstream elements which device entered the location at this URI (either cid-URL or location-by-reference URI).

9.3. SIP signaling requirements for proxy servers

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SP-33 SIP Proxy servers processing emergency calls:

1. If the proxy interprets dial plans on behalf of user agents, the proxy MUST look for the local emergency dial string at the location of the end device and MAY look for the home dial string. If it finds it, the proxy MUST:
 - *Insert a Geolocation header field. Location-by-reference MUST be used because proxies must not insert bodies.
 - *Include the Geolocation "inserted-by" and "used-for-routing" parameters with its own hostname (which should match the Via it inserts) on the inserted-by.
 - *Map the location to a PSAP URI using LoST.
 - *Add a Route header with the PSAP URI.
 - *Replace the Request-URI (which was the dial string) with the service URN appropriate for the emergency dial string.
 - *Route the call using normal SIP routing mechanisms.

2. If the proxy recognizes the service URN in the Request URI, and does not find a Route header, it MUST query a LoST server. If a location was provided (which should be the case), the proxy uses that location to query LoST. The proxy may have to dereference a location by reference to get a value. If a location is not present, and the proxy can query a LIS which has the location of the UA it MUST do so. If no location is present, and the proxy does not have access to a LIS which could provide location, the proxy MUST supply a default location (See [Section 6.11 \(Default location\)](#)). The location (in the signaling, obtained from a LIS, or default) MUST be used in a query to LoST with the service URN received with the call. The resulting URI MUST be placed in a Route header added to the call.
3. The proxy SHOULD NOT modify any parameters in Geolocation header fields received in the call. It MAY add a Geolocation header field. Such an additional location SHOULD NOT be used for routing; the location provided by the UA should be used.
4. Either a P-Asserted-Identity [\[RFC3325\] \(Jennings, C., Peterson, J., and M. Watson, "Private Extensions to the Session Initiation Protocol \(SIP\) for Asserted Identity within Trusted Networks," November 2002.\)](#) or an Identity header field [\[RFC4474\] \(Peterson, J. and C. Jennings, "Enhancements for Authenticated Identity Management in the Session Initiation Protocol \(SIP\)," August 2006.\)](#), or both, SHOULD be included to identify the sender. For services which must support emergency calls from unauthenticated devices, valid identity may not be available. Proxies encountering a P-Asserted-Identity will need to pass the header to the PSAP, which is in a different domain. [\[RFC3325\] \(Jennings, C., Peterson, J., and M. Watson, "Private Extensions to the Session Initiation Protocol \(SIP\) for Asserted Identity within Trusted Networks," November 2002.\)](#) requires a "spec(T)" to determine what happens if the "id" privacy service, or a Privacy header is present and requests privacy. In the absence of another spec(T), such proxies should pass the header unmodified if and only if the connection between the proxy and the PSAP is, as far as the proxy can determine, protected by TLS with mutual authentication using keys reliably known by the parties, encrypted with no less strength than AES and the local regulations governing the PSAP do not otherwise specify.

10. Call backs

ED-64/SP-34 Devices device SHOULD have a globally routable URI in a Contact: header field which remains valid for several minutes past the time the original call containing the URI completes unless the device registration expires and is not renewed.

SP-35 Call backs to the Contact: header URI received within 30 minutes of an emergency call must reach the device regardless of call features or services that would normally cause the call to be routed to some other entity.

SP-36 Devices MUST have a persistent AOR URI either in a P-Asserted-Identity header field or From protected by an Identity header field suitable for returning a call some time after the original call. Such a call back would not necessarily reach the device that originally placed the call.

11. Mid-call behavior

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ED-65/SP-37 During the course of an emergency call, devices and proxies MUST initiate a call transfer upon receipt of REFER request within the dialog with method=INVITE and the Referred-by: header field [\[RFC3515\]](#) (Sparks, R., "The Session Initiation Protocol (SIP) Refer Method," April 2003.) in that request.

12. Call termination

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ED-66 There can be a case where the session signaling path is lost, and the user agent does not receive the BYE. If the call is hung up, and the session timer (if implemented) expires, the call MAY be declared lost. If in the interval, an incoming call is received from the domain of the PSAP, the device MUST drop the old call and alert for the (new) incoming call. Dropping of the old call MUST only occur if the user is attempting to hang up; the domain of an incoming call can only be determined from the From header, which is not reliable, and could be spoofed. Dropping an active call by a new call with a spoofed From header field would be a DoS attack.

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13. Disabling of features

ED-67/SP-38 User Agents and proxies MUST disable features that will interrupt an ongoing emergency call, such as:

- *Call Waiting
- *Call Transfer
- *Three Way Call
- *Hold
- *Outbound Call Blocking

when an emergency call is established. Also see ED-74 in [Section 14 \(Media\)](#).

ED-68/SP-39 The emergency dial strings SHOULD NOT be permitted in Call Forward numbers or speed dial lists.

ED-69/SP-40 The User Agent and Proxies MUST disable call features which would interfere with the ability of call backs from the PSAP to be completed such as:

- *Do Not Disturb
- *Call Forward (all kinds)

ED-70 Call backs SHOULD be determined by retaining the domain of the PSAP which answers an outgoing emergency call and instantiating a timer which starts when the call is terminated. If a call is received from the same domain and within the timer period, sent to the Contact: or AoR used in the emergency call, it should be assumed to be a call back. The suggested timer period is 5 minutes. [\[RFC4916\] \(Elwell, J., "Connected Identity in the Session Initiation Protocol \(SIP\)," June 2007.\)](#) may be used by the PSAP to inform the UA of the domain of the PSAP. Recognizing a call back from the domain of the PSAP will not always work, and further standardization will be required to give the UA the ability to recognize a call back.

14. Media

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ED-71 Endpoints MUST send and receive media streams on RTP [\[RFC3550\] \(Schulzrinne, H., Casner, S., Frederick, R., and V. Jacobson, "RTP: A Transport Protocol for Real-Time Applications," July 2003.\)](#).

ED-72 Normal SIP offer/answer [\[RFC3264\] \(Rosenberg, J. and H. Schulzrinne, "An Offer/Answer Model with Session Description Protocol](#)

[\(SDP\)," June 2002.\)](#) negotiations MUST be used to agree on the media streams to be used.

ED-73 Endpoints supporting voice MUST support G.711 A law (and mu Law if they are intended be used in North America) encoded voice as described in [\[RFC3551\] \(Schulzrinne, H. and S. Casner, "RTP Profile for Audio and Video Conferences with Minimal Control," July 2003.\)](#). It is desirable to include wideband codecs such as AMR-WB in the offer.

ED-74 Silence suppression (Voice Activity Detection methods) MUST NOT be used on emergency calls. PSAP call takers sometimes get information on what is happening in the background to determine how to process the call.

ED-75 Endpoints supporting Instant Messaging (IM) MUST support both [\[RFC3428\] \(Campbell, B., Rosenberg, J., Schulzrinne, H., Huitema, C., and D. Gurle, "Session Initiation Protocol \(SIP\) Extension for Instant Messaging," December 2002.\)](#) and [\[RFC4975\] \(Campbell, B., Mahy, R., and C. Jennings, "The Message Session Relay Protocol \(MSRP\)," September 2007.\)](#).

ED-76 Endpoints supporting real-time text MUST use [\[RFC4103\] \(Hellstrom, G. and P. Jones, "RTP Payload for Text Conversation," June 2005.\)](#). The expectations for emergency service support for the real-time text medium are described in [\[RFC5194\] \(van Wijk, A. and G. Gybels, "Framework for Real-Time Text over IP Using the Session Initiation Protocol \(SIP\)," June 2008.\)](#), Section 7.1.

ED-77 Endpoints supporting video MUST support H.264 per [\[I-D.ietf-avt-rtp-rfc3984bis\] \(Wang, Y., Even, R., Kristensen, T., and R. Jesup, "RTP Payload Format for H.264 Video," October 2010.\)](#).

15. Testing

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ED-78 INVITE requests to a service URN ending in ".test" indicates a request for an automated test. For example, "urn:service.sos.fire.test". As in standard SIP, a 200 (OK) response indicates that the address was recognized and a 404 (Not found) that it was not. A 486 (Busy Here) MUST be returned if the test service is busy, and a 404 (Not found) MUST be returned if the PSAP does not support the test mechanism.

ED-79 In its response to the test, the PSAP MAY include a text body (text/plain) indicating the identity of the PSAP, the requested service, and the location reported with the call. For the latter, the PSAP SHOULD return location-by-value even if the original location delivered with the test was by-reference. If the location-by-reference was supplied, and the dereference requires credentials, the PSAP SHOULD use credentials supplied by the LIS for test purposes. This alerts the LIS that the dereference is not for an actual emergency call and location hiding techniques, if they are being used, may be employed for

this dereference. Use of SIPS for the request would assure the response containing the location is kept private

ED-80 A PSAP accepting a test call SHOULD accept a media loopback test [\[I-D.ietf-mmusic-media-loopback\]](#) (Sivachelvan, C., Venna, N., Jones, P., Stratton, N., Roychowdhury, A., and K. Hedayat, "An Extension to the Session Description Protocol (SDP) for Media Loopback," July 2010.) and SHOULD support the "rtp-pkt-loopback" and "rtp-start-loopback" options. The user agent would specify a loopback attribute of "loopback-source", the PSAP being the mirror. User Agents should expect the PSAP to loop back no more than 3 packets of each media type accepted (which limits the duration of the test), after which the PSAP would normally send BYE.

ED-81 User agents SHOULD perform a full call test, including media loopback, after a disconnect and subsequent change in IP address not due to a reboot. After an initial test, a full test SHOULD be repeated approximately every 30 days with a random interval.

ED-82 User agents MUST NOT place a test call immediately after booting. If the IP address changes after booting, the UA should wait a random amount of time (in perhaps a 30 minute period, sufficient for any avalanche restart to complete) and then test.

ED-83 PSAPs MAY refuse repeated requests for test from the same device in a short period of time. Any refusal is signaled with a 486 or 488 response.

16. Security Considerations

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Security considerations for emergency calling have been documented in [\[RFC5069\]](#) (Taylor, T., Tschofenig, H., Schulzrinne, H., and M. Shanmugam, "Security Threats and Requirements for Emergency Call Marking and Mapping," January 2008.), and [\[I-D.ietf-geopriv-arch\]](#) (Barnes, R., Lepinski, M., Cooper, A., Morris, J., Tschofenig, H., and H. Schulzrinne, "An Architecture for Location and Location Privacy in Internet Applications," October 2010.).

17. IANA Considerations

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This document has no actions for IANA.

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